IN THE CLAIMS:

This listing of claims will replace all prior versions and listings of claims in the

application.

1.   (currently amended) A digital signal processing system comprising:

a)   a signal processor adapted to be connected to a pair of signal transmission paths

and through which signal commands on one of the paths are transmitted; and

b)   a signal canceller operatively associated with the signal processor to subtract

~~undesired~~ echo signals from received signals on the other of the paths, the signal canceller

employing Auxiliary-Vector filtering.

2.   (original) The system according to claim 1, wherein the signal canceller contains

an array of filter coefficients and wherein the Auxiliary Vector filtering and multiplies the

signals by the array of filter coefficients and subtracts the result from the transmitted signal

commands.

3.   (currently amended) The system according to claim ~~1~~ 2, wherein the signal

canceller is adaptive in being capable of changing the filter coefficients while receiving data

from the transmitted signal commands and from the received signals.

4.   (original) The system according to claim 2, wherein the filter coefficients are

obtained using the algorithm:

$$R=Y^T*Y$$

$$V=Y^T*X$$

$$G=(R*V-(V^T*R*V)*V)/\text{norm}(R*V-(V^T*R*V)*V)$$

$$U=(G^T*R*V)/(G^T*R*G)$$

$$w=V-U*G$$

where the vector Y contains N values of the received signal y[n] where N is the number of filter

coefficients, X contains one value x[n] of the transmitted signal commands, G is the Auxiliary

Vector, U is a scalar which minimizes output variances of the filter coefficients and w is a vector

containing the filter coefficients.

5.   (original) The system according to claim 1, wherein voice signals are processed.

6.   (currently amended) The system according to claim 1, wherein the signal

processor is the digital signal processor of a voice mail system, and wherein the signal

commands include voice mail prompts ~~and the undesired signals include echo~~.

7.   (currently amended) A digital signal processing system comprising:

a)   a near end and a far end connected by a pair of signal transmission paths;

b)   a signal processor at the near end through which signal commands are transmitted

from the near end to the far end;

c)   a signal transducing device at the far end to receive the signal commands and to

transmit to the near end signals indicating the state of the signal transducing device; and

d)   a signal canceller operatively associated with the signal processor to subtract ~~undesired~~ echo signals from the received signals, the signal canceller employing Auxiliary-Vector filtering.

8.   (original) The system according to claim 7, wherein the signal canceller contains an array of filter coefficients and wherein the Auxiliary Vector filtering takes the signals transmitted from the far end and multiplies the transmitted signals by the array of filter coefficients and subtracts the result from the signal commands transmitted from the near end.

9.   (currently amended) The system according to claim ~~1~~ 8, wherein the signal canceller is adaptive in being capable of changing the filter coefficients while receiving data from the signal commands transmitted from the near end and the signals transmitted from the far end.

10.   (original) The system according to claim 8, wherein the filter coefficients are obtained using the algorithm:

$$R=Y^T*Y$$

$$V=Y^T*X$$

$$G=(R*V-(V^T*R*V)*V)/\text{norm}(R*V-(V^T*R*V)*V)$$

$$U=(G^T*R*V)/(G^T*R*G)$$

w=V-U*G

where the vector Y contains N values of the received signal y[n] where N is the number of filter

coefficients, X contains one value x[n] of the transmitted signal commands, G is the Auxiliary

Vector, U is a scalar which minimizes output variances of the filter coefficients and w is a vector

containing the filter coefficients.

11.   (original) The system according to claim 7, wherein voice signals are processed.

12.   (currently amended) The system according to claim 7, wherein the signal

processor is the digital signal processor of a voice mail system, the signal transducing device is a

telephone set, wherein the signal commands include voice mail prompts and the undesired

signals include echo.

13.   (currently amended) A digital signal processing method comprising:

a)   transmitting signal commands from a near end including a signal processor over a

first signal path to a far end including a signal transducing device;

b)   receiving over a second signal path from the far end to the near end signals

indicating the state of the signal transducing device; and

c)   cancelling undesired echo signals from the signals received at the near end

utilizing a signal canceller employing Auxiliary-Vector filtering.

14. (currently amended) The method according to claim ~~7~~ 13, wherein the ~~signal~~ cancelling includes providing an array of filter coefficients and wherein the Auxiliary Vector filtering takes the signals transmitted from the far end and multiplies the transmitted signals by the array of filter coefficients and subtracts the result from the signal commands transmitted from the near end.

15. (currently amended) The method according to claim ~~7~~ 14, wherein the ~~signal~~ cancelling is adaptive in being capable of changing the filter coefficients while receiving data from the signal commands transmitted from the near end and the signals transmitted from the far end.

16. (original) The method according to claim 14, wherein the filter coefficients are obtained using the algorithm:

$$R=Y^T*Y$$

$$V=Y^T*X$$

$$G=(R*V-(V^T*R*V)*V)/norm(R*V-(V^T*R*V)*V)$$

$$U=(G^T*R*V)/(G^T*R*G)$$

$$w=V-U*G$$

where the vector Y contains N values of the received signal y[n] where N is the number of filter coefficients, X contains one value x[n] of the transmitted signal commands, G is the Auxiliary Vector, U is a scalar which minimizes output variances of the filter coefficients and w is a vector

containing the filter coefficients.

17. (currently amended) The method according to claim ~~7~~ 13, wherein voice signals are processed.

18. (currently amended) The method according to claim ~~7~~ 13, wherein the signal processor is the digital signal processor of a voice mail system, the signal transducing device is a telephone set, wherein the signal commands include voice mail prompts ~~and the undesired signals include echo~~.

19. (currently amended) A program storage device readable by a machine embodying a program of instructions executable by the machine for signal processing in which undesired signals are cancelled, the instructions comprising:

a) transmitting signal commands from a near end including a signal processor over a first signal path to a far end including a signal transducing device;

b) receiving over a second signal path from the far end to the near end signals indicating the state of the signal transducing device; and

c) cancelling ~~undesired~~ echo signals from the signals received at the near end utilizing a signal canceller employing Auxiliary-Vector filtering.

· 20. (currently amended) The program storage device according to claim 19, wherein

the ~~signal~~ cancelling includes providing an array of filter coefficients and wherein the Auxiliary Vector filtering takes the signals transmitted from the far end and multiplies the transmitted signals by the array of filter coefficients and subtracts the result from the signal commands transmitted from the near end.

21.   (currently amended) The program storage device according to claim ~~19~~ 20, wherein the ~~signal~~ cancelling is adaptive in being capable of changing the filter coefficients while receiving data from the signal commands transmitted from the near end and the signals transmitted from the far end.

22.   (original) The program storage device according to claim 20, wherein the filter coefficients are obtained using the algorithm:

$R=Y^T*Y$

$V=Y^T*X$

$G=(R*V-(V^T*R*V)*V)/\text{norm}(R*V-(V^T*R*V)*V)$

$U=(G^T*R*V)/(G^T*R*G)$

$w=V-U*G$

where the vector Y contains N values of the received signal y[n] where N is the number of filter coefficients, X contains one value x[n] of the transmitted signal commands, G is the Auxiliary Vector, U is a scalar which minimizes output variances of the filter coefficients and w is a vector containing the filter coefficients.

23. (original) The program storage device according to claim 19, wherein voice signals are processed.

24. (currently amended) The program storage device according to claim 19, wherein the signal processor is the digital signal processor of a voice mail system, the signal transducing device is a telephone set, wherein the signal commands include voice mail prompts ~~and the undesired signals include echo~~.